

Sipura SPA-2000 Manual Configuration Instructions



Only settings that deviate from factory defaults are included in this document. If your ATA was previously connected to a different service, you will need to use the screenshots to verify that all settings are correct; alternatively, you may reset the ATA to factory defaults (recommended), as described in section 4.

1. Plug in the ATA

Connect a telephone handset, network, and power to the ATA, and wait briefly while it discovers its IP address via DHCP. For more information on connecting your ATA, or to set a Static IP Address manually, please consult your User Manual.

2. Determine the ATA's IP Address

To find out the ATA's IP Address, dial "****", then "110#". Note down the IP Address spoken out and enter it into the address bar of a Web browser.

Example: If the address is "10.34.238.6", enter this as "<http://10.24.238.6/>".

If the address read out is "0.0.0.0", the ATA was unsuccessful in acquiring an IP address via DHCP -- contact your network administrator before proceeding.

3. Reset the ATA to Factory Defaults (optional).

Note: This will clear all custom settings from the ATA; if you wish to retain these for later reference, please make a note of them (for example, by printing the configuration pages, or saving a copy of them to disk).

To perform the reset, dial "****" on the attached telephone handset, followed by "73738#", then "1" to confirm; then hang up. Once the phone is hung up, the ATA will restart with its settings reset to factory defaults.

4. Configure the ATA via its web interface

Bring up the ATA's Web configuration page, as described in section 2 of this document. Click on "Admin Login", then "advanced". Clicking on each of the System, SIP, Regional, Line 1 and (optionally) Line 2 tabs in turn, make the following changes; click on the "Submit All Changes" button when done with each page:

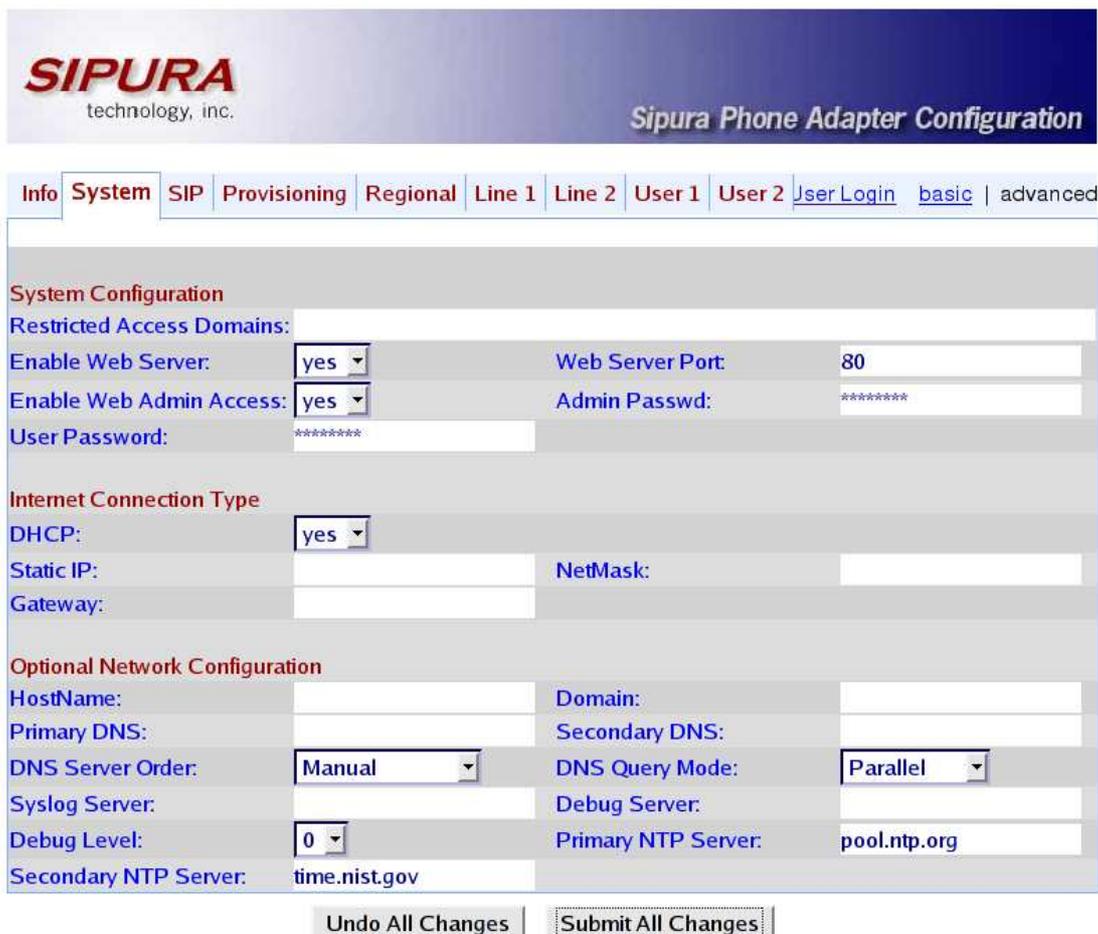
System

System Configuration

Admin Passwd: <choose a secure password for administration>
User Password: <may be the same as the Admin Passwd>

Optional Network Configuration

Primary NTP Server: pool.ntp.org
Secondary NTP Server: time.nist.gov



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Sipura Phone Adapter Configuration

Info System SIP Provisioning Regional Line 1 Line 2 User 1 User 2 User Login basic advanced

System Configuration

Restricted Access Domains:

Enable Web Server: Web Server Port:

Enable Web Admin Access: Admin Passwd:

User Password:

Internet Connection Type

DHCP:

Static IP: NetMask:

Gateway:

Optional Network Configuration

HostName: Domain:

Primary DNS: Secondary DNS:

DNS Server Order: DNS Query Mode:

Syslog Server: Debug Server:

Debug Level: Primary NTP Server:

Secondary NTP Server:

SIP

NAT Support Parameters

Handle VIA received: yes
Handle VIA rport: yes
Insert VIA received: yes
Insert VIA rport: yes
STUN Enable: yes
STUN Server: stun.freshtel.net

NAT Support Parameters			
Handle VIA received:	<input type="text" value="yes"/>	Handle VIA rport:	<input type="text" value="yes"/>
Insert VIA received:	<input type="text" value="yes"/>	Insert VIA rport:	<input type="text" value="yes"/>
Substitute VIA Addr:	<input type="text" value="no"/>	Send Resp To Src Port:	<input type="text" value="no"/>
STUN Enable:	<input type="text" value="yes"/>	STUN Test Enable:	<input type="text" value="no"/>
STUN Server:	stun.freshtel.net	EXT IP:	
EXT RTP Port Min:		NAT Keep Alive Intvl:	15

Regional (Australia only)

Call Progress Tones

Dial Tone: 400@-19,425@-19;30(*0/1+2)
Busy Tone: [425@-10;10\(.4/4/1\)](#)
Reorder Tone: [425@-10;10\(.2/2/1\)](#)
Ring Back Tone: [400@-19,425@-19,450@-19;*\(.4/2/1+2+3,.4/2/1+2+3,0/2/0\)](#)

SIPURA technology, inc. Sipura Phone Adapter Configuration

Info	System	SIP	Provisioning	Regional	Line 1	Line 2	User 1	User 2	User Login	basic	advanced
Call Progress Tones											
Dial Tone:	400@-19,425@-19;30(*0/1+2)										
Second Dial Tone:	420@-19,520@-19;10(*0/1+2)										
Outside Dial Tone:	420@-16;10(*0/1)										
Prompt Tone:	520@-19,620@-19;10(*0/1+2)										
Busy Tone:	425@-10;10(.4/4/1)										
Reorder Tone:	425@-10;10(.2/2/1)										
Off Hook Warning Tone:	480@-10,620@0;10(.125/125/1+2)										
Ring Back Tone:	400@-19,425@-19,450@-19;*(.4/2/1+2+3,.4/2/1+2+3,0/2/0)										
Confirm Tone:	600@-16;1(.25/25/1)										
SIT1 Tone:	985@-16,1428@-16,1777@-16;20(.380/0/1,.380/0/2,.380/0/3,0/4/0)										
SIT2 Tone:	914@-16,1371@-16,1777@-16;20(.274/0/1,.274/0/2,.380/0/3,0/4/0)										
SIT3 Tone:	914@-16,1371@-16,1777@-16;20(.380/0/1,.380/0/2,.380/0/3,0/4/0)										
SIT4 Tone:	985@-16,1371@-16,1777@-16;20(.380/0/1,.274/0/2,.380/0/3,0/4/0)										
MWI Dial Tone:	350@-19,440@-19;2(.1/1/1+2);10(*0/1+2)										
Cfwd Dial Tone:	350@-19,440@-19;2(.2/2/1+2);10(*0/1+2)										
Holding Tone:	600@-19;*(.1/1/1,.1/1/1,.1/9.5/1)										
Conference Tone:	350@-19;20(.1/1/1,.1/9.7/1)										
Secure Call Indication Tone:	397@-19,507@-19;15(0/2/0,.2/1/1,.1/2.1/2)										

Regional (all countries)

Miscellaneous

Time Zone:

GMT+10:00 <or as appropriate>

FXS Port Impedance:

600	<North America>
270+750 150nF	<most of Europe>
220+820 120nF	<Australia, New Zealand>
220+820 115nF	<Austria, Bulgaria, Germany, Slovakia, South Africa>
370+620 310nF	<UK, India>

Miscellaneous			
Set Local Date (mm/dd):		Set Local Time (HH/mm):	
Time Zone:	GMT+10:00	FXS Port Impedance:	220+820 120nF
FXS Port Input Gain:	-3	FXS Port Output Gain:	-3
DTMF Playback Level:	-16	DTMF Playback Length:	.1
Detect ABCD:	yes	Playback ABCD:	yes
Caller ID Method:	Bellcore(N.Amer,China)	FXS Port Power Limit:	3

Undo All Changes

Submit All Changes

Line 1 (and 2, if using a different Firefly number on the second line)

NAT Mapping Enable: yes
NAT Keep Alive Enable: yes
NAT Keep Alive Msg:

Proxy and Registration

Proxy: sip.freshtel.net
Make Call Without Reg: yes
Ans Call Without Reg: yes

Subscriber Information

Display Name: <your full name>
User ID: <your Firefly number>
Password: <your Firefly password>


Sipura Phone Adapter Configuration

Info
System
SIP
Provisioning
Regional
Line 1
Line 2
User 1
User 2
User Login
basic
advanced

Line Enable:	<input type="text" value="yes"/>	SAS Enable:	<input type="text" value="no"/>
MOH Server:	<input type="text"/>	SAS DLG Refresh Intvl:	<input type="text" value="30"/>
NAT Mapping Enable:	<input type="text" value="yes"/>	SAS Inbound RTP Sink:	<input type="text"/>
SIP Port:	<input type="text" value="5060"/>	NAT Keep Alive Enable:	<input type="text" value="yes"/>
EXT SIP Port:	<input type="text"/>	NAT Keep Alive Msg:	<input type="text"/>
SIP TOS/DiffServ Value:	<input type="text" value="0x68"/>	NAT Keep Alive Dest:	<input type="text" value="\$PROXY"/>
RTP TOS/DiffServ Value:	<input type="text" value="0xb8"/>	SIP Debug Option:	<input type="text" value="none"/>
Network Jitter Level:	<input type="text" value="high"/>	SIP 100REL Enable:	<input type="text" value="no"/>
Blind Attn-Xfer Enable:	<input type="text" value="no"/>	SIP Proxy-Require:	<input type="text"/>
Auth Resync-Reboot:	<input type="text" value="yes"/>	SIP Remote-Party-ID:	<input type="text" value="no"/>

Proxy and Registration

Proxy:	<input type="text" value="sip.freshtel.net"/>	Use Outbound Proxy:	<input type="text" value="no"/>
Outbound Proxy:	<input type="text"/>	Use OB Proxy In Dialog:	<input type="text" value="yes"/>
Register:	<input type="text" value="yes"/>	Make Call Without Reg:	<input type="text" value="yes"/>
Register Expires:	<input type="text" value="3600"/>	Ans Call Without Reg:	<input type="text" value="yes"/>
Use DNS SRV:	<input type="text" value="no"/>	DNS SRV Auto Prefix:	<input type="text" value="no"/>
Proxy Fallback Intvl:	<input type="text" value="3600"/>		

Subscriber Information

Display Name:	<input type="text" value="John Smith"/>	User ID:	<input type="text" value="80001234"/>
Password:	<input type="text" value="*****"/>	Use Auth ID:	<input type="text" value="no"/>
Auth ID:	<input type="text"/>		
Mini Certificate:	<input type="text"/>		
SRTP Private Key:	<input type="text"/>		

Audio Configuration

Preferred Codec: G729a

NOTE: It is vital that the dial plans are entered correctly – as there is not enough space on a page to show them as one line, you should **copy and paste** one of the below dial plans to ensure it is correct. Missing a space or getting a character wrong will stop them from working.

Dial Plan (Australia only -- user has no local dialing prefix set)

```
(*123|*xxxxxxx|<000:61000> S0|<112:61112> S0|<106:61106> S0|61000 S0|61112
S0|61106 S0|0[2378][2-
6]xxxxxx|0[4569]xxxxxx|611300xxxx|611800xxxx|6113xxxx|6118xxxx|[1-9]x.)
```

Dial Plan (Australia only -- user has local dialing prefix set)

```
(*123|*xxxxxxx|000 S0|112 S0|106 S0|001[0-9]x.|0[4569]xxxxxx|0[2378][2-9]xxxxxx|[2-
9]xxxxxx|1300xxxx|1800xxxx|13xxxx|18xxxx|1[4-79]x.)
```

Audio Configuration			
Preferred Codec:	G729a	Silence Supp Enable:	no
Use Pref Codec Only:	no	Echo Canc Enable:	yes
G729a Enable:	yes	Echo Canc Adapt Enable:	yes
G723 Enable:	yes	Echo Supp Enable:	yes
G726-16 Enable:	yes	FAX CED Detect Enable:	yes
G726-24 Enable:	yes	FAX CNG Detect Enable:	yes
G726-32 Enable:	yes	FAX Passthru Codec:	G711u
G726-40 Enable:	yes	FAX Codec Symmetric:	yes
DTMF Tx Method:	Auto	FAX Passthru Method:	NSE
Hook Flash Tx Method:	None	FAX Process NSE:	yes
Release Unused Codec:	yes		
Dial Plan			
Dial Plan:	(*123 *xxxxxxx 000 S0 112 S0 106 S0 001[0-9]x. 0[4569]xxxxxx 0[2378][2-9]x.)		
Enable IP Dialing:	no		
FXS Port Polarity Configuration			
Idle Polarity:	Forward	Caller Conn Polarity:	Forward
Callee Conn Polarity:	Forward		

Undo All Changes

Submit All Changes

Dial Plan (UK only -- user has no local dialing prefix set)

```
(*123|*xxxxxxx|<999:44999> S0|<112:44999> S0|44999 S0|44112 S0|4418000
S0|4418001999 S0|440800112999 S0|0[2-9]xxxxxx|[1-9]x.)
```

Dial Plan (UK only -- user has local dialing prefix set)

(*123|*xxxxxxx|999 S0|112 S0|18000 S0|18001999 S0|0800112999 S0|[2-9]xxxxxx|[2-9]xxxxxx|[2-9]xxxx|[2-9]xxxx|[1237]xxxxxxxx|01xxxxxxxx|0[045689]x.l1x.)

Dial Plan (US and Canada) <no change>

Once these changes are made and the "Submit All Changes" button is clicked, your ATA will reboot, and you should be ready to place and receive calls with Freshtel.

Dialling:

To dial a Firefly number or voicemail, prefix the number with a star, e.g. ***80001234**, ***123**.

If you have configured a local dialling prefix for your account via the [Freshtel website](#), you can dial local, national, mobile, and international PSTN numbers exactly as you would from a land line phone in your area; e.g. **08007654321** (US), **9557745** (US – local), **0061399156415** (UK – international), **99156415** (Australia – local).

If you have not configured a local dialling prefix, numbers are dialled in full (including country code but without any international dial prefix), e.g. **61399156415**, **18007654321**. Land lines and mobiles in Australia can be dialled with an initial **0**, e.g. **0399156415**.